

## LWA Signal Delays and Time Tagging

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For a beam-forming channel of an LWA station, consider a plane wavefront arriving from the direction of the beam. We can imagine that the sky contains only a point source in that direction. Let  $t_0$  be the wavefront's time of arrival at the station's reference position (an arbitrarily chosen place, but probably near the center of the station's antennas for convenience). The same wavefront induces a signal in each of the antenna stands, and those signals propagate through the telescope system. Ultimately they are combined, and we end up with one polarization pair of output signals for that beam. (From now on in this memo, we ignore the fact that there is a pair of signals and consider just one of them.)

Each antenna's signal suffers various time delays before and after combining until it finally arrives at the output. The total delay can be represented as

$$T = G_i + C_i + g_i + c_i + x + L + F(B) \quad (1)$$

where  $G_i$  is the geometrical delay in propagation from the reference position to antenna  $i$  (possibly negative);  $C_i$  is the delay from antenna  $i$  to the corresponding digitizer (which I will call the cable delay, even though it also includes propagation through the front end and analog receiver electronics);  $g_i$  and  $c_i$  are compensating delays inserted in the digital electronics, nominally equal to  $-G_i$  and  $-C_i$ , respectively;  $x > 0$  is a fixed extra delay inserted in the digital electronics in order to keep  $x + c_i + g_i$  positive for all antennas and all possible beam directions;  $L$  is the extra signal latency caused by pipelining of the digital processing; and  $F(B)$  is the delay through the final filter (after combining), which is a function of the selected filter bandwidth  $B$ . Here we have neglected the fact that some of the delays vary with frequency across the bandwidth due to the frequency responses of the components, on the assumption that care was taken to minimize such effects<sup>2</sup>.

The sum of the first four terms in (1) is nominally zero for all antennas, and this is the mechanism by which the array is made to have high gain in the desired beam direction. Sometimes it may deliberately be made non-zero for some antennas in order to accomplish beam shaping, but we regard these as intentional departures from nominal. Delay  $x$  is implemented separately for each antenna's signal (along with  $g_i$  and  $c_i$ ), but its value is made exactly the same for all. Delay  $L$  is distributed through the processing path of each signal, and it would naturally be different for each antenna because the individual signals are added to the sum at different places along a chain, but the design has carefully included a compensating delay in each antenna's path in such a way that the net delay at

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<sup>2</sup> Dispersion in the antenna and cable, if not well matched among the antennas, may be partly cancelled in the digital filter that forms part of  $g_i$ . Dispersion common to all antennas may be partly cancelled in the final filter. Otherwise, digital filters will be designed to have linear phase and hence be dispersionless. In general there will be some residual dispersion, so that the compensating delays  $g_i$  and  $c_i$  are accurate only at one or a few frequencies in the processed bandwidth.

the final summing point is the same for all. The final filter with delay  $F(B)$  is applied only to the combined signal, so it is naturally the same for all antennas.

All of the delays implemented in the DP subsystem are known to it, and therefore can be taken into account. In particular, the nominal delay  $T_{\text{nom}} = x + L + F(B)$  is known. This does not mean that  $T_{\text{nom}}$  is always the same for the same selected bandwidth because the latency  $L$  depends partly on FIFO buffers whose state is not directly under control of the designers. Thus  $L$  could change across a power cycle, system reset, or major re-configuration. It will nevertheless be measured within the hardware so that its value during any observation is known<sup>3</sup>.

Based on the above discussion, we propose to consider the time of a signal sample that arrives at the output of the final filter when the station clock time is  $t_1$  to be

$$t_{\text{tag}} = t_1 - T_{\text{nom}} ,$$

since this is our best estimate of  $t_0$ , the time of arrival of the corresponding wavefront at the reference position. To the extent that the DP subsystem provides a "time tag" for any sample, this is the value that it should use. Alternatively, if the timing of samples is the responsibility of the Data Aggregation and Communication Subsystem, DP can provide it with the current value of  $T_{\text{nom}}$  via the Monitor/Control Subsystem. In this way, the peculiar latency  $L$  of the DP subsystem, along with its somewhat arbitrary value of  $x$ , need not be taken into account by later processors, nor need they account for variations of delay with bandwidth  $B$ . This is especially important when combining the station signals at a correlator because it is possible that one or more components of  $T_{\text{nom}}$  will not be the same for all stations. There will still be a peculiar delay for each station due to differences between  $T$  and  $T_{\text{nom}}$  (for example, the cable delays  $C_i$  may not all be absolutely compensated, even if their *differences* within the station have been accurately removed by  $c_i$ ), and this will need to be calibrated astronomically. If residual dispersion in the station beam signals is significant and varies among stations, the calibration may need to be a function of frequency.

Various copies of the station clock may exist in different hardware modules and in computers. All should be synchronized (but perhaps to varying accuracies) via distributed reference signals and messages. The module responsible for labeling the times of output samples must contain a particularly accurate copy of the station clock, and to that extent it might be considered the "master" copy. The DP2 module should contain a counter driven by the 196 MHz sampling signal, and this counter's value can serve as a station clock with resolution 5 nsec and ambiguity interval determined by the counter's length (where 44 bits is sufficient for 24 hours).

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<sup>3</sup> There are several ways to do this, including intermittent direct measurement and the use of periodic time markers in the signals. The exact implementation has not yet been decided.